

MUSIC IV PROGRAMMER'S MANUAL

by

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The authors would like to express their appreciation to Mrs. C. C. Lochbaum for her work in separating BE FAP from BE SYS 5 for use with the Music IV program and the write-up of Section 4.

Section 1. Introduction.

This material is intended to be a handbook and guide to the use of the Music IV program, which is the most recent version of the computer music programs developed at Bell Telephone Laboratories. The material includes descriptive parts intended to introduce the reader to the music routines, cookbook procedures for using the program, and detailed technical descriptions of the programs. The technical descriptions are intended to be sufficiently complete so that a competent programmer can adapt the music for programs for use at any IBM 7090-7094 installation and can modify the programs to meet his own particular musical objectives. This handbook is intended to be accompanied by a tape containing card images of all the music programs and another memorandum describing the operation of the BE FAP compiler. These materials should be sufficient to introduce the Music IV program at another computer installation.

We suggest that the potential user familiarize himself in a general way with the material in this handbook before making a detailed study of the technical parts. The various sections contain the following materials:

Section 2 - This section is an over-all description of the operation of the music program, which might be given in a technical paper. It does not go into sufficient detail to allow the use of the program but provides a good introduction.

Section 3 - This section gives a description of the tape containing the computer music programs and outlines how to get started in adapting these programs to a computer installation.

Section 4 - This section gives the details of the BE FAP compiler. This program is essential for compiling orchestras, and adapting it for use at another computer installation is probably the most difficult programming job required by the music programs.

Section 5 - This section is a cookbook outline of how to execute a music program including compiling an orchestra and playing a score with the orchestra.

Section 6 - This section contains detailed technical descriptions of the various music programs.

Section 2. Generation of music by the Music IV program.

The objective of the Music IV program is to produce a sequence of numbers which can be converted to sound by means of a digital-to-analog converter and a smoothing filter. This process is schematized in Figure 1. It has been described in detail in other publications ("The Digital Computer as a Musical Instrument" by M. V. Mathews, Science, Vol. 142, No. 3592, pp.553-557, Nov. 1, 1963) and little more will be said about it here. The digital-to-analog converter, the smoothing filter and the loudspeaker by which the numerical samples are converted to sound are pieces of physical equipment which must be obtained for any computer installation desiring to reproduce sound from numbers. Likewise a suitable output program must be written to transmit the numbers from the computer memory into the converter. The details of this program depend completely upon the particular computer installation and analog-to-digital converter which are used. Consequently, the output program must be written by the computer user and is not included in this group of music programs.

In the music program the sequence of numbers corresponding to musical sounds are generated by simulated instruments built up from combinations of unit generators. Each unit generator is a small block of computer instructions performing a given operation such as that of an oscillator, an adder, or a random noise generator. Figure 2 shows a diagram of a typical simulated instrument. It produces periodic tones in which the wave shape can be controlled, the attack and decay characteristics can be controlled, and frequency can be varied by means of both a periodic and a random vibrato. The instrument specifications in terms of the coding necessary to specify ten of these instruments is indicated at the bottom of the figure. The first line gives the name of the instrument, WAIL; the next six lines specify the unit generators in the instrument. Each unit generator is specified by its type, for example, OSCIL, standing for oscillator. The interconnectivity of the unit generators is specified by giving the origin of the inputs to each unit generator. The first oscillator U1, for example, has inputs P4 and C3 which are supplied by the composer on a note card at the beginning of each note to be played by the instrument. Oscillator U5, by contrast, has as inputs the output of oscillator U1 and adder U4. The computer description of the unit generators is completed by specifying any other parameters which they may require. For example, the oscillators refer to functions F1 through F20 which are stored in the computer memory and which determine the wave shape of the oscillation.

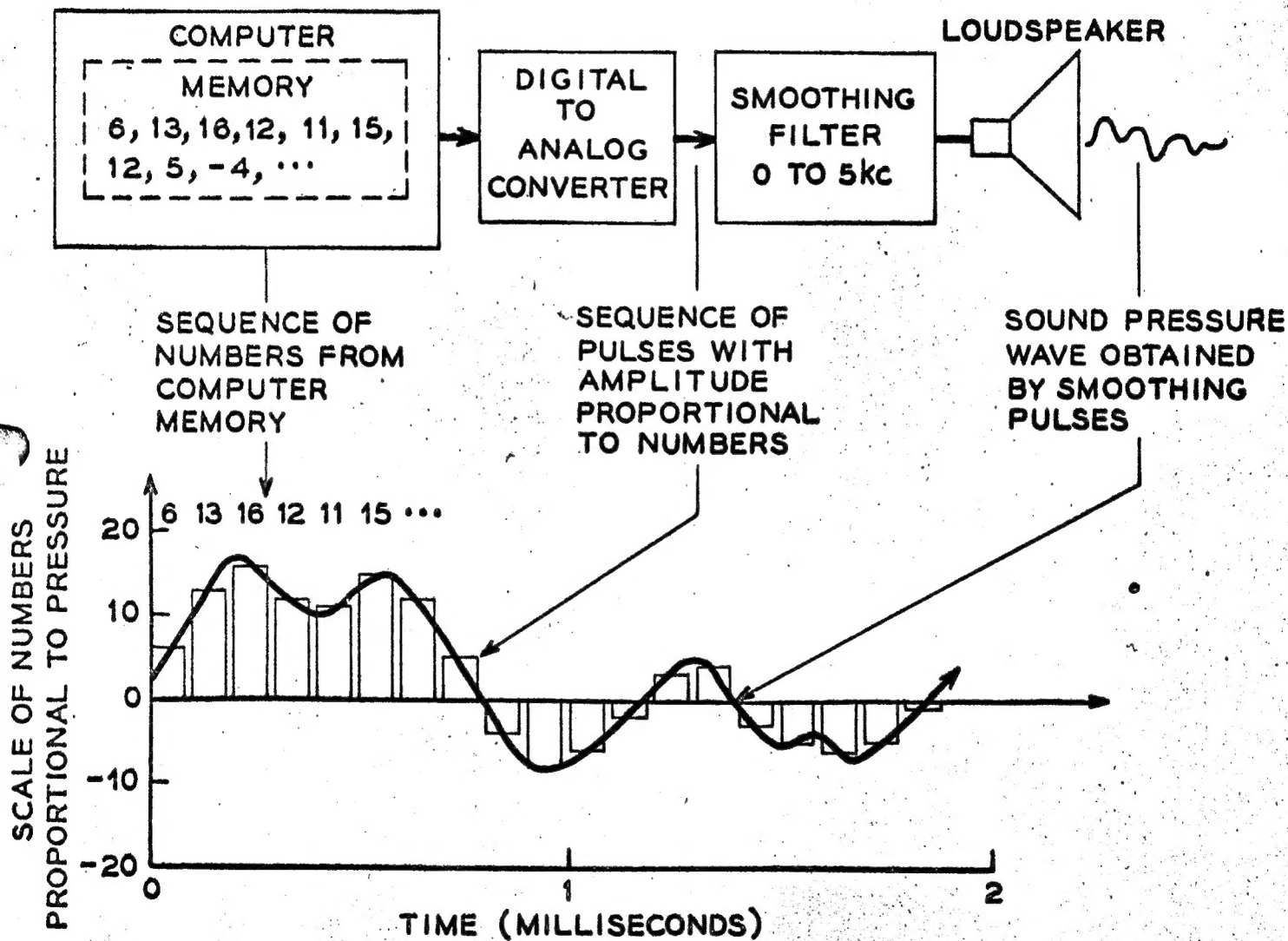
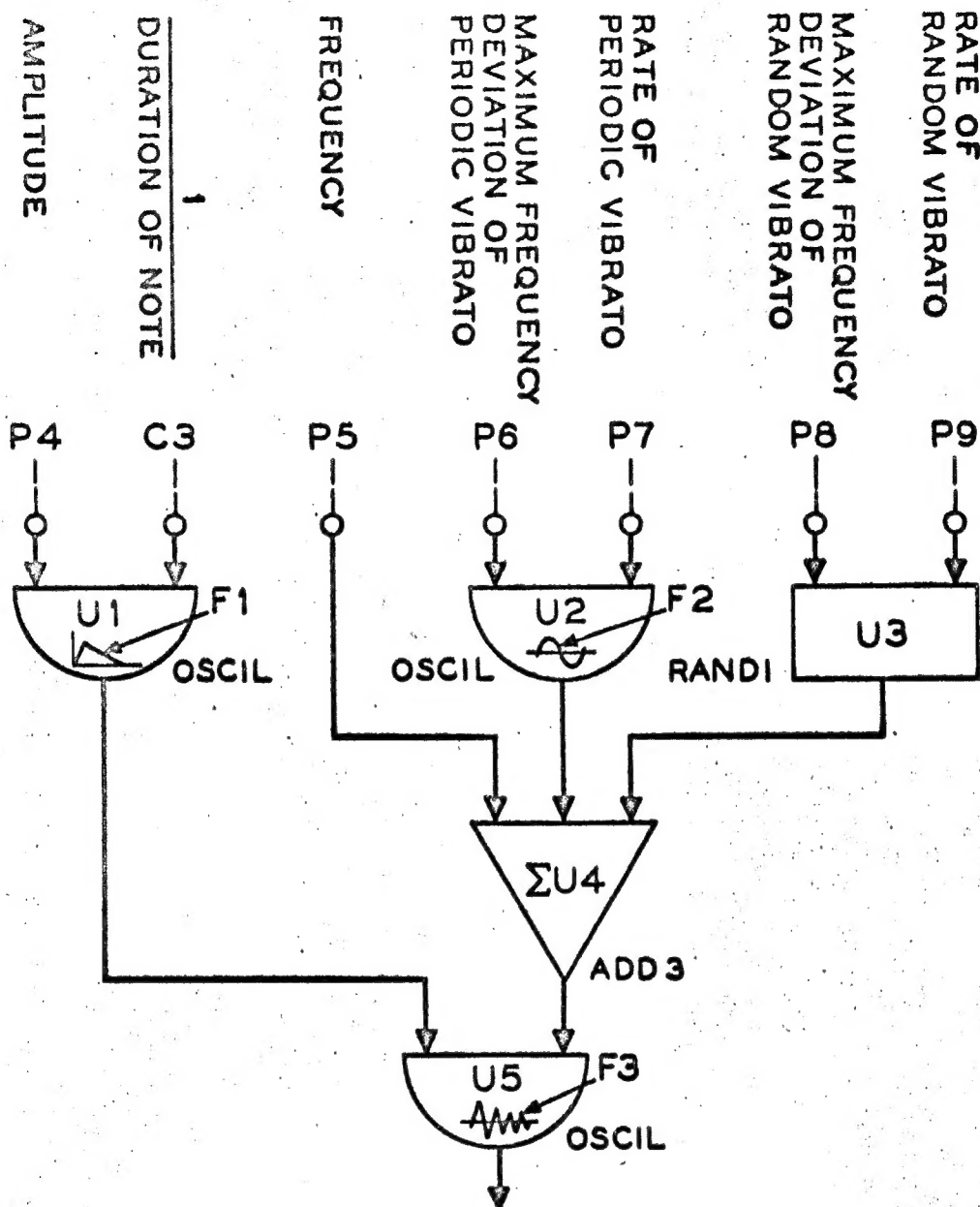


Fig 1
Matthews

BLOCK DIAGRAM AND COMPUTER CODE FOR 10 WAILS



WAIL	INSTR	
	ØSCIL	P4, C3, F1
	ØSCIL	P6, P7, F2
	RANDI	P8, P9
	ADD3	P5, U2, U3
	ØSCIL	U1, U4, F3
	ØUT	U5
	END	
WAIL	CØUNT	10
	FINE	

Fig 2

NOTE CARDS

OP	P1	P2	P3	P4	P5	P6	P7	P8	P9	P10	P11	P12
PLA	2	0	0.75	1	466	0	0	7.0	6			
PLA	3	0.5	0.50	2	116	1.7	6	0	0			

UNUSED
 RANDOM VIBRATO BANDWIDTH (CPS)
 RANDOM VIBRATO AMPLITUDE (CPS)
 PERIODIC VIBRATO FREQUENCY (CPS)
 PERIODIC VIBRATO AMPLITUDE (CPS)
 FREQUENCY (CPS)
 LOUDNESS (ARBITRARY SCALE)
 DURATION (BEATS)
 STARTING TIME (BEATS)
 INSTRUMENT NO.

The computer instructions resulting from the instrument specifications are produced by the computer by executing BE FAP, a compiling program. In this compilation there are many self-checking features which detect most types of errors the composer may make. For example, if he lists either too many or too few parameters for a given unit generator or puts down the wrong kind of parameter, say a function instead of an input parameter, the compiler will inform the composer of his error and refuse to proceed. A successful compilation results in an orchestra program.

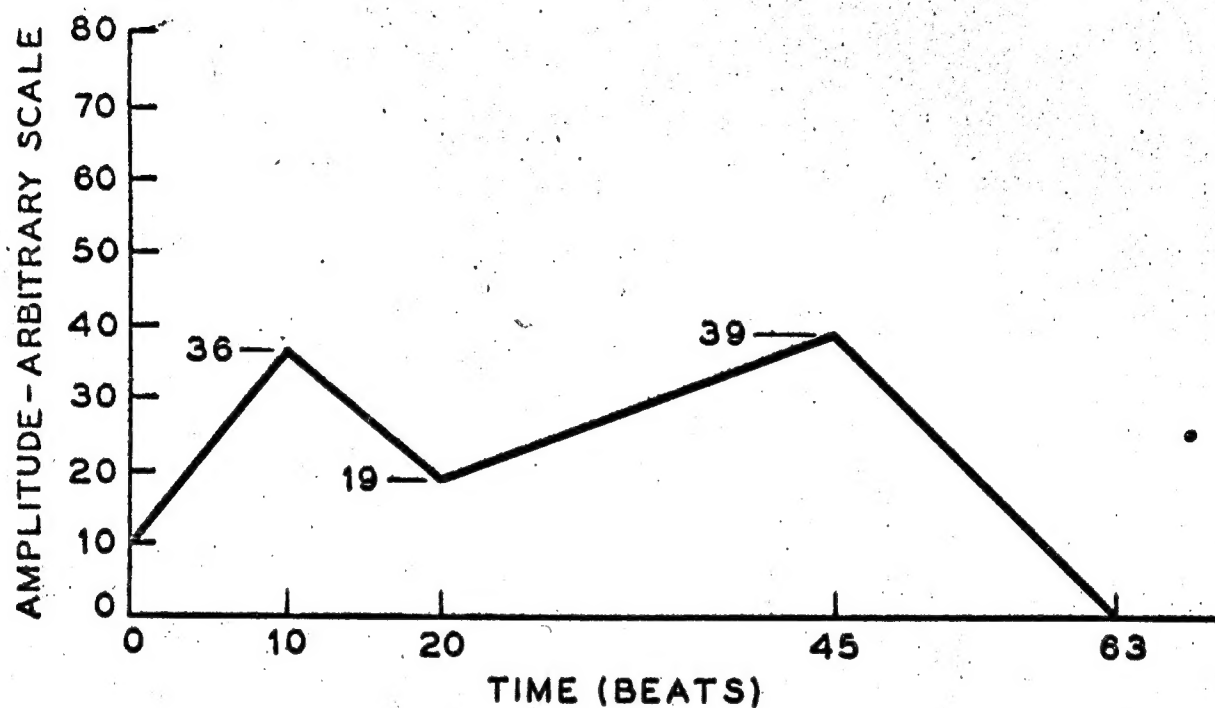
Ten instruments all having the same form are compiled by the single specification WAIL COUNT 10. The last line FINE indicates that there are no additional instruments in the orchestra.

After the orchestra has been compiled, a score consisting of notes to be played by the instruments is "played" by means of a set of playing programs which operate together with the orchestra program. Notes to be played by the instruments can be generated by the computer using automatic composing features of the playing program or the notes can be punched manually, one note per computer card. In this case each card has room for an operation code and 12 numerical parameters as shown on Figure 3. Here two notes are played on the WAIL instruments #2 and #3. The instrument number is given in P1, the starting time in P2, and the duration in P3. Both starting time and duration are given in beats and the beats are converted into seconds by an arbitrary velocity transformation. Loudness in P4 is given on an arbitrary scale. Note frequency is given in P5 in cycles per second. The amplitudes and rates of the periodic and random vibratos are given in P6 and P9. On this particular instrument, P10 through P12 are not used. In more complicated instruments they may be used.

In addition to specifying notes the computer cards may be used to specify compositional functions. These are functions which change over the course of the composition and are typically used to control parameters such as tempo or loudness. The format for such functions is indicated on Figure 4. They are built up from line segments. On the computer card the time and amplitude of the end point of each line segment is specified. Many very short line segments can be used to describe rapidly changing portions of a function and a few long line segments can be used for the slowly changing portions. Thus an efficient coding is achieved which requires a minimum of writing on the part of the composer.

The operation of the playing program is schematized on Figure 5. The program has been divided into three parts, Pass I, Pass II, and Pass III, to facilitate the use of the computer as a

COMPOSITIONAL FUNCTION



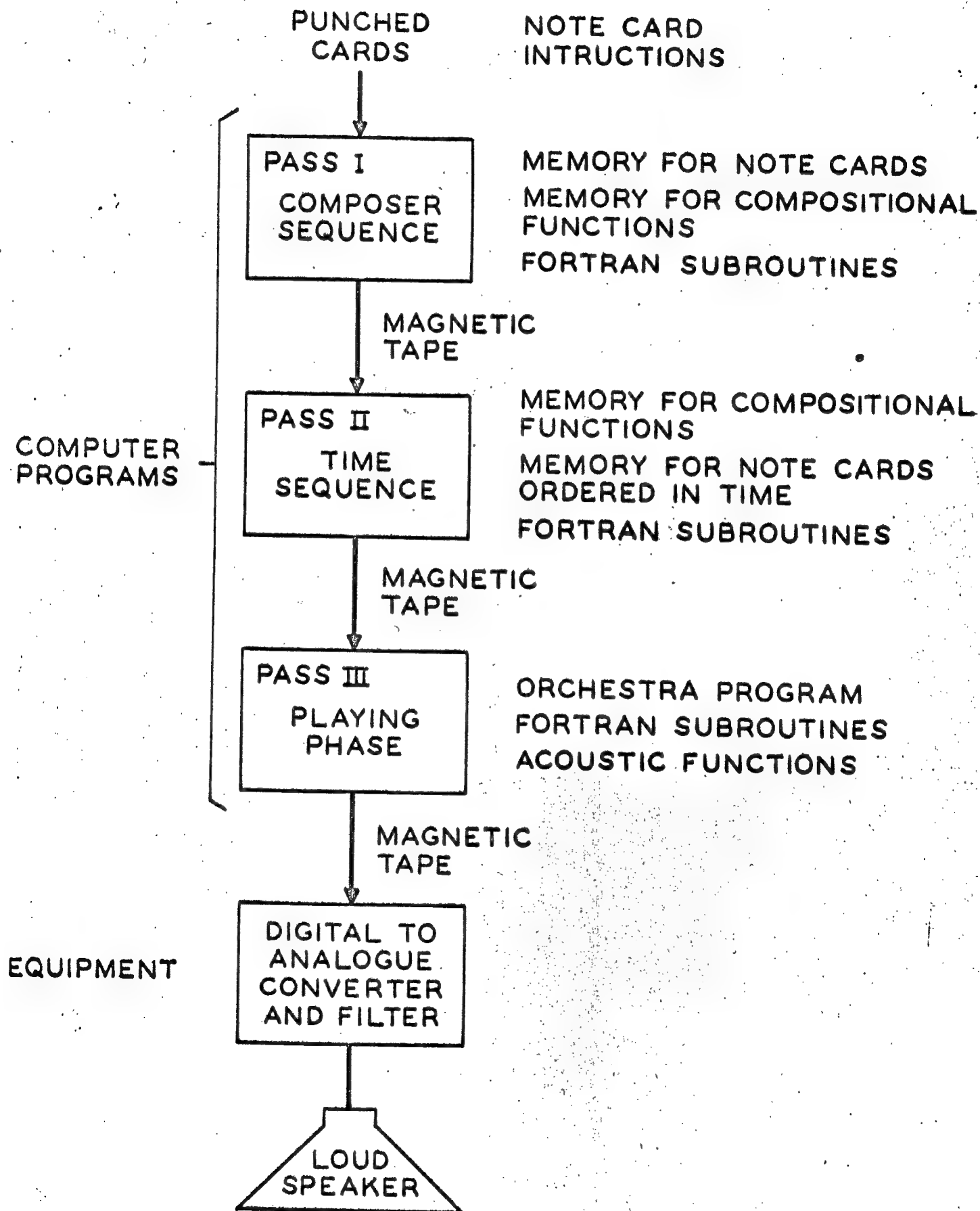
OP	P1	P2	P3	P4	P5	P6	P7	P8	P9	P10	P11	P12
110	0	0	0	10	10	36	20	19	45	39	63	0

composing medium. The input to Pass I consists of the cards which the composer prepares. These are note cards such as were shown on Figure 3 plus instruction cards which specify how the note cards will be modified before they are played. Pass I contains a memory for 800 note cards thus allowing sequences of notes to be replayed with or without modification. The same memory also may be used to store compositional functions which are involved in modifying the note cards. The actual modifications are performed by FORTRAN subroutines. FORTRAN has proven to be a very satisfactory language in which to write many of the music programs.

In Pass I the cards are processed in the sequence in which the composer has inserted them into the computer. In Pass II the notes are sorted into the time sequence in which they will be played. Consequently, in Pass II it is convenient to apply transformations which are functions of time such as, for example, altering the tempo.

Pass III is the playing phase in which the notes specified directly by the composer or generated by the computer are presented to the orchestra program. The output is shown being directly converted to sound. It is more likely that the numerical samples will be stored on a magnetic tape or disk for subsequent conversion to sound. Also the figure shows communication between the various Passes by means of digital magnetic tapes. If a disk is available it will do these jobs better. Details such as these can and indeed must be altered by the particular user of the program to fit his computer installation.

MUSIC IV PROGRAM



Section 5. Operating procedure.

It is assumed at this point that the following preparations have been made:

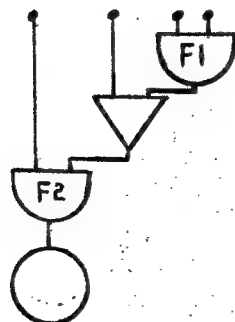
1. The absolute BINARY program of BE FAP (File 1) has been used to assemble the CRUNCH deck of BE FAP (File 2) and the BE FAP compiler is available for use (See Section 4);
2. Files 3 and 4 have been punched, and the FAP routines DREAD and SORT have been assembled with BE FAP and the FORTRAN routines of File 4 have been compiled.

It remains (1) to design, define and compile with BE FAP an orchestra of the composer's choice and (2) to run the three passes of the music program. We shall discuss first the

Preparation of an orchestra

As has been indicated earlier, the orchestra is designed by the composer. The orchestra compiler is a CRUNCH deck of BE FAP coding (given in part 1 File 3 of tape) and the composer writes additional cards defining the instruments in the orchestra of his choice. The CRUNCH deck together with the cards defining the orchestra are then treated as one source deck to be compiled by BE FAP. The resulting object deck, which is to be run in Pass III is thus tailored to the requirements of the composer.

The first step in constructing the orchestra is to design the instruments. An example of an instrument (WAIL) has already been given in Section 2. We shall consider here an instrument of even simpler type: one which emits some specified wave shape at any amplitude and frequency where the vibrato of the note can be controlled. The instrument type should be designed in block diagram form where each block is a unit generator. For example, the diagram of the above instrument is as follows:



unit 1 oscillator

unit 2 adder

unit 3 oscillator

unit 4 output box

Oscillators are unit generators which have three parameters. The first two are inputs, which, when read from left to right on the diagram specify the amplitude and frequency respectively of the wave shape emitted by the oscillator. The function number of this wave shape is designated by the third parameter. Adders put out the sum of the input signals. The output box is required to be the last unit in the design of any instrument. In this instrument a function F1 put out by unit 1 is added to a frequency specification for unit 3 in the adder (unit 2). Therefore, the wave form, F2, emitted by unit 3 is frequency modulated by unit 1. Inputs to unit 1 and the left inputs to units 2 and 3 come from note cards in the score, whereas the right inputs to units 2 and 3 and the input to unit 4 come from units 1, 2, and 3 respectively.

After an instrument type has been designed it remains to define the type for the computer. It is necessary first to name the instrument. Any name of up to six alphanumeric characters may be chosen. The first card in the definition contains this name in the location field (cols. 1-6) and INSTR in the operation field (cols. 8-12). The last card is an END card. Between these two is a sequence of cards - one for each of the unit generators in the design. The order of these cards determines the interconnectivity of the unit generators, that is, units which serve as inputs to a given unit should be listed ahead of the given unit. The source of all inputs must be specified for each unit generator in the instrument, and these specifications are listed in the variable field (column 16). The functions, if any, required by the unit are listed immediately after the inputs. Consequently, the definition of the instrument designed above is given by cards 3 through 8 in Table 1.

The list of available unit generators is given in Table 2, together with specifications, diagrams, and descriptions. In writing cards for an instrument definition, the inputs to a unit generator should be listed in the order given by reading from left to right on the diagram. The designation of inputs are of four types:

1. P_n - for those inputs coming directly from the nth field of note cards in the score.
2. C_n - for those inputs coming indirectly from note cards via converting functions where n is the number of the converting function.
3. U_n - for those inputs coming from some other unit generator where n is the position of this other unit in the list of units defining the instrument.
4. K_n - for those inputs which are fixed constants where n refers to the nth number in a list of constants. A list of

numerical quantities (in which decimal points are required) may be entered into the orchestra deck by writing the operation code `CØNSTS` and a list of decimal numbers separated by commas and enclosed in parentheses in the variable field. Such a card must precede all instrument definitions for which there are any unit generators using constant inputs.

Functions associated with units are of two types:

1. F_n - those which are fixed throughout the performance where n is the number of the function.
2. P_n - those which vary in the course of the performance where the function number to be used is given in the n^{th} field of a note card in the score.

A stored function is requested and numbered by a GEN data card read in Pass I. The function is actually computed or generated in Pass III for use in the playing phase of Pass III. Details on generating subroutines, converting functions, and fields for parameters on note cards can be found in Section 6.

To construct an orchestra after having defined the instrument types it is necessary merely to call for the instruments by name in the desired order. A single instrument can be obtained by one card with the name of the type in the operation field. Several instruments of the same type can be called for on a single card by writing the name in the location field, the code `CØUNT` in the operation field and the desired number of instruments in the variable field.

The score refers to the instruments by number and thus this sequence of cards calling for instrument types must be ordered to correspond to this numbering. If in the score, instrument numbers have been skipped, then positions in the list of instrument names should be marked by cards having `DUMMY` in their operation fields. There should be one `DUMMY` card for each instrument omitted. (The `CØUNT` code is not applicable for `DUMMY` instruments.)

Either monophonic or two-channel stereophonic digital output signals can be produced by the orchestra. The former requires no specification and the latter may be obtained by preceding all instrument definitions by a card having the operation code `STEREØ`. If this mode is chosen, then the `ØUT` units of all instrument types should have a second input which specifies the proportion (0 to 1) of input to channel A (1 minus this proportion is used for channel B). If the orchestra is defined in `STEREØ` and the second input is omitted from the `ØUT` boxes, then the channels will be balanced. The digital output signal is multiplexed, that is, the samples of the generated sound are computed for the two channels alternately.

Finally, all definitions and calls for instruments are followed by a card which has FINE in its operation field. Therefore, an orchestra using three instruments as defined above, which are numbered 1, 2, and 5 in the score and which are to play at constant amplitude and produce a balanced stereophonic output would be called for by the cards given in Table 1. To compile the orchestra thus defined these cards written by the composer should be placed directly behind the orchestra CRUNCH deck and the totality should thus be compiled by BE FAP. The resulting object program as a binary card deck may be quite large, and, therefore, tape or disk storage is recommended for the binary card images. (See Section 6 - Orchestra program for advice on the first compilation.)

To run the Music IV program

The Music IV program in its entirety is comprised of three separate programs, referred to as Passes.

Pass I - The deck includes PASSI (main program), DREAD, WRITER, and CON plus any PLF subroutines of the composer's choice (See Section 6). Input data for this pass is the score, that is, the composition in the form of note cards prepared by the composer, plus additional cards which may call upon subroutines. These cards are processed in the order in which they are read and an intermediate tape, A4, is generated as output of Pass I. (Tape A4 is, of course, specifically associated with only the BTL version, and other systems may use different tapes or disk storage.) This tape contains all the note card images written directly by the composer together with any note card images generated by PLF subroutine. The note card images on A4 can occur in any order relative to their final order of playing.

Pass II - The deck includes PASSII (main program), SORT, and CON. The intermediate tape A4 from Pass I is read by PASSII and the note card images are sorted into the sequence in which the notes will be played in Pass III. A velocity scaling to introduce changes in tempo is applied. The adjusted, time-ordered note card parameters are then written onto a second intermediate tape, B2 (again specific to BTL).

Pass III - The deck includes PASSIII (main program), the orchestra deck, the generating routines GEN07, GEN09 and/or others supplied by the composer, convert functions of the composer's choice, any necessary mathematical routines such as square root or trigonometric functions, and an output program for the samples of the generated sound. This routine is not supplied since its operation will depend upon the installation where used. Its requirements are described in Section 6. The intermediate tape B2 of Pass II containing the sorted note card

images is read by the master program and the orchestra program is called upon to generate samples of the acoustic output. These are written on tape A5 by the output routine. (A5 is again specific to BTL.)

Therefore, the sequence of cards that the user should prepare is as follows:

To be compiled by BE FAP	{ Orchestra CRUNCH deck Instrument definitions (last card - FINE)
To be loaded and run	{ PASSI (main) DREAD WRITER CON (PLF _n - optional routines supplied by composer)
Data to be read by Pass I	{ Data - e.g., GEN cards and score
To be loaded and run	{ PASSII (main) SORT CON
To be loaded and run	{ PASSIII (main) - Orchestra program - from compilation above Output program - written by user { CVTon - converting functions - used as needed } { GENon - generating routines - " " " }

As an aid to understanding this sequence of operations we include a printout from a compilation and run at BTL. See Figure 6. Checks in the left-most column denote BE SYS 5 control cards. Much of the printing pertaining to subroutine entry points and origins results from the BE SYS 5 loader. Other outputs are from the Music IV routines and show in Pass I, the note cards for the score and in Pass II, the sorted score. Pass III has no particular printed output except for an indication of sections and a final comment from the output routine. The listing has been annotated and hopefully will be self-explanatory. Other relevant details may be found in the descriptions of the individual programs given in Section 6.

Fig. 6

✓ J08 AAL3,05 JEM BLDG 2-5 12 HRS 34 MIN 15 SEC 05/11/64
INDECK FAP

PAGE 1 PASS 1 COMMENTS AND ALTER CARD LISTING

12:34:15 05/11/64 J08 AAL3

CRUNCH / /

PAGE 2

12:34:15 05/11/64 J08 AAL3

SUBROUTINE ENTRY POINTS

INITL	26301
GEN	26315
INSERT	26354
PLAY	26444
REMOVE	26516
TER	26563

TRANSFER VECTOR

00000	236563000060	CVT00
00001	236563000160	CVT01
00002	236563000260	CVT02
00003	236563000360	CVT03
00004	236563000460	CVT04
00005	236563000560	CVT05
00006	236563000660	CVT06
00007	236563000760	CVT07
00010	236563000860	CVT08
00055	236563040560	CVT09
00056	236563040660	CVT46
00057	236563040760	CVT47
00060	236563041060	CVT48
00061	272545000060	GEN00
00062	272545000160	GEN01
00063	272545000260	GEN02
00064	272545000360	GEN03
00065	272545000460	GEN04
00066	272545000560	GEN05
00067	272545000660	GEN06
00070	272545000760	GEN07
00071	272545001060	GEN08
00072	272545001160	GEN09
00073	272545010060	GEN10
00074	272545010160	GEN11
00075	272545010260	GEN12
00076	272545010360	GEN13
00077	272545010460	GEN14
00100	272545010560	GEN15
00101	272545010660	GEN16
00102	272545010760	GEN17
00103	272545011060	GEN18
00104	272545011160	GEN19
00105	272545020060	GEN20
00106	624746646360	SPBUT
00107	624725452460	SPEND
00110	624746646301	SPBUT1

provision for
49 converting functions

21 generating routines

output routines

1093

*** ORCHESTRA ***

c7 c2 c9 c2

BELLS INSIR

OSCIL C3,C2,P5
 OSCIL C7,C2,P11
 OSCIL C9,C2,P12
 OSCIL U2,U3,P7

ADD2 C5,U4

OSCIL U1,U5,P9

OUT U6

END

BELLS COUNT 2

INSTRUMENT 1

PZE ..012+1

PZE ..014+1

PZE ..016+1

PZE ..018+1

PZE ..024+1

PZE ..028+1

PZE ..035+1

INSTRUMENT 2

PZE ..043+1

PZE ..045+1

PZE ..047+1

PZE ..049+1

PZE ..055+1

PZE ..059+1

PZE ..068+1

FINE

Addresses of Outputs
for Unit Generators

.06

26602 0 00000 0 26653

26603 0 00000 0 26703

26604 0 00000 0 26733

26605 0 00000 0 26763

26606 0 00000 0 27011

26607 0 00000 0 27026

26610 0 00000 0 27054

26611 0 00000 0 27071

26612 0 00000 0 27121

26613 0 00000 0 27151

26614 0 00000 0 27201

26615 0 00000 0 27227

26616 0 00000 0 27244

26617 0 00000 0 27272

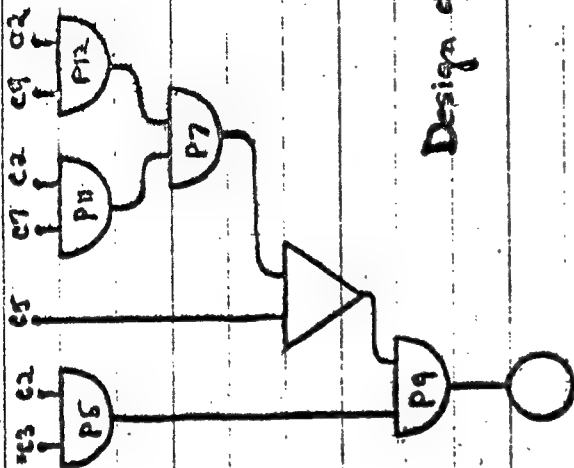
27441

LITERALS

27442 IS THE FIRST LOCATION NOT USED BY THIS PROGRAM

NO ERROR IN ABOVE ASSEMBLY

Design of Instrument "Beus"



tape specifications

Note: Checks in leftmost column indicate BE SYSTEM control cards. Crosses on right indicate lines of print produced by BE SYSTEM. All other lines result from MUSIC II.

✓ SCR MYTAPE A4
 ✓ SCR MYTAPE B2
 ✓ OUTPUT MYTAPE A5
 ✓ LOADS A5
 ✓ TRA

PLF1 NOT IN DECK
 PLF2 NOT IN DECK
 PLF3 NOT IN DECK
 PLF4 NOT IN DECK
 PLF5 NOT IN DECK
 PLF6 NOT IN DECK
 PLF7 NOT IN DECK
 PLF8 NOT IN DECK
 PLF9 NOT IN DECK
 PLF10 NOT IN DECK
 PLF11 NOT IN DECK
 PLF12 NOT IN DECK
 PLF13 NOT IN DECK
 PLF14 NOT IN DECK
 PLF15 NOT IN DECK

SUBROUTINE ENTRY LOCATIONS
 000000 00126 DREAD
 (SPH) 77014 (FIL)
 X 00640 WRITER 01325 CBN
 77024 ENDJOB 77000 (RWT)
 77026 (WLR) 77031 X

SUBROUTINE ORIGINS
 000000 00100
 00640 00640
 WRITER 01314
 CBN 01435

Pass I

PROTECTED BUFFERS HAVE BEEN SET UP FROM 10000 TO

GEN 09	1	0	1	512	0
GEN 07	2	0	50	412	0
GEN 07	3	0	20	216	0
ETC	0	0	0	0	0
GEN 07	4	0	256	256	0
GEN 07	5	0	512	0	0
1	0.00	0.20	0.20	0.20	0.20
1	0.25	0.20	0.20	0.20	0.20
1	0.50	0.40	0.20	0.20	0.20
1	1.00	0.20	0.20	0.20	0.20
1	1.25	0.20	0.20	0.20	0.20
1	1.50	0.40	0.20	0.20	0.20
1	2.00	0.20	0.20	0.20	0.20
1	2.25	0.20	0.20	0.20	0.20
1	2.50	0.30	0.20	0.20	0.20
1	2.87	0.10	0.20	0.20	0.20
1	3.00	0.90	0.20	0.20	0.20
2	0	0.20	0.20	0.20	0.20
2	0.25	0.20	0.20	0.20	0.20
2	0.50	0.40	0.20	0.20	0.20
2	1.00	0.20	0.20	0.20	0.20
2	1.25	0.40	0.20	0.20	0.20
2	1.50	0.20	0.20	0.20	0.20
2	2.00	0.20	0.20	0.20	0.20
2	2.25	0.20	0.20	0.20	0.20
2	2.50	0.30	0.20	0.20	0.20
2	2.87	0.10	0.20	0.20	0.20
2	3.00	0.90	0.20	0.20	0.20

Data for Pass I

Note: Encircled numbers were not punched on data cards but result from carry feature of DREAD.

TER

to load in compiled orchestra

[illegible]

PROTECTED BUFFERS HAVE BEEN SET UP FROM 40000 TO

831
SEC.

END OUTPUT FILE 0 SAMPLES OUT OF RANGE

DATE	TIMES:	ON	OFF	ELAPSED	SETUP	0	CHARGED	1401	SYSTEM	5	F0RTRAN	0	FAP	1	RUN	3	DUMPS	PIT	P0T	C0DE	X	
05/11	AAL3	12.5708	12.6000	0.0292	0.	0.0292	0.0172	0.0136	0.	0.0105	0.0050	0.0001	6431	3482	448C	4						

PRINTED ON 1401 SYSTEM 3

Compilation time (orchestra) run time

Table 1
Example of an Orchestra Definition

Cards	Cols 1 Location field	8 Operation field	16 Variable field
1		STEREO	
2		CONSTS	(5.)
3	WOBBLE	INSTR	
4		OSCIL	P6, P7, F1
5		ADD2	C4, U1
6		OSCIL	K1, U2, F2
7		OUT	U3
8		END	
9	WOBBLE	COUNT	2
10		DUMMY	
11		DUMMY	
12		WOBBLE	
13		FINE	

Table 2

Unit Generator Specifications

Note: The inputs to the unit generators are described in general by I_n where n indicates the number of the input counting from left to right on the diagram. Functions, if any, follow the inputs and are denoted by G . The subscript i used in the description of the units denotes sample number.

1. $\emptyset UT \ I1$



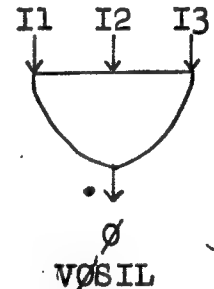
This unit generator must be the last unit in all instruments. It adds the sample specified by $I1$ to the final acoustical output.

In the STEREO mode, $\emptyset UT$ may have two inputs $\emptyset UT \ I1, I2$ in which $I2$ (between 0 and 1) specifies the proportion of the instrument's output going to channel A and $1-I2$ specifying the proportion of the output going to channel B. (If $I2$ is omitted, half the output goes to both channels.)

2. $\emptyset SCIL \ I1, I2, G$

3. $C\emptyset SIL \ I1, I2, G$

4. $V\emptyset SCIL \ I1, I2, I3$



These generate functions and oscillations according to the equations

$$\left. \begin{array}{l} \emptyset SCIL \\ C\emptyset SIL \end{array} \right\} \emptyset_1 = I1_1 * G([S_1] \bmod 512)$$

$$S_{1+1} = S_1 + I2_1$$

In $\emptyset SCIL$, S_0 is set to zero at the beginning of each note.
In $C\emptyset SIL$ it is not.

When these units are used as oscillators, $I1$ specifies the amplitude of the oscillation and $10000 \times I2 / 512$ is the frequency.

Table 2 (Page 2)

For VØSCIL

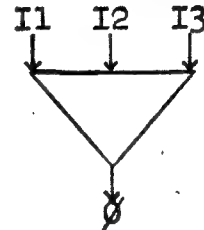
$$\phi_1 = I1_1 * I3 ([S_1] \bmod 512),$$

thus the function number is considered as an input and can be changed during the course of the note. If $I3_1$ is not an integer, it is truncated to the next smallest integer.

5. ADD2 I1, I2

6. ADD3 I1, I2, I3

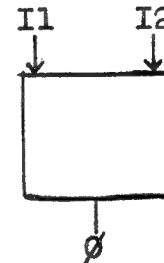
7. ADD4 I1, I2, I3, I4



These units add. Thus for example in ADD3

$$\phi_1 = I1_1 + I2_1 + I3_1$$

8. RANDI I1, I2



This generator produces a function

$$\phi_1 = I1_1 \times \text{Random Function}(I2_1)$$

where Random Function($I2_1$) is a low pass random function having a bandwidth of approximately $5000 \times I2 / 512$ cps and an amplitude ranging from -1 to +1. The function is formed by connecting independent random numbers with line segments. The number of samples on each line segment is $512 / I2$.

9. RANDH I1, I2

This generator produces a function

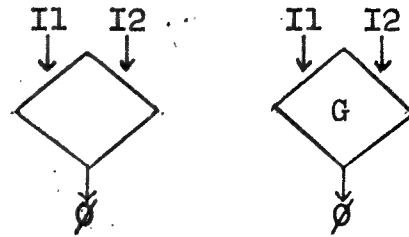
$$\phi_1 = I1_1 \times \text{Random Number}(I2_1)$$

where Random Number($I2_1$) is a succession of independent random numbers that change every $512 / I2$ sample. In other words this generator holds each random number for $512 / I2$ samples.

Table 2 (Page 3)

10. MULT I1,I2

11. VFMULT I1,I2,G

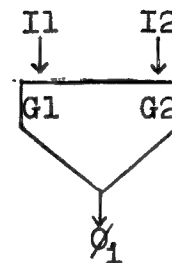


These multipliers perform according to the equations

$$\text{MULT} \quad \phi_1 = I1_1 * I2_1$$

$$\text{VFMULT} \quad \phi_1 = I1_1 * G(I2_1)$$

12. RESØN I1,I2,G1,G2



This generator produces oscillations which are the product of function scanned at a fixed rate times a function scanned at a variable rate according to the equation

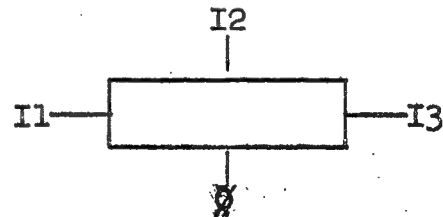
$$\phi_1 = I1_1 * G1([S_1] \text{ mod } 512) * G2([T_1] \text{ mod } 512)$$

$$S_{i+1} = S_i + 1$$

$$T_{i+1} = T_1 + I2_1$$

The function G1 can be used to introduce fixed formant frequencies. The function G2 is usually of a type $1/2(1 - \cos 2\pi x / 512)$ which goes to zero at $x=1$ and $x=512$ and thus smoothes any discontinuities in F_1 .

13. FILTER I1,I2,I3



This unit simulates a single pole pair bandpass filter with the difference equation

$$\phi_1 = I2_1 + I1_1 \phi_{1-1} - I3_1 \phi_{1-2}$$

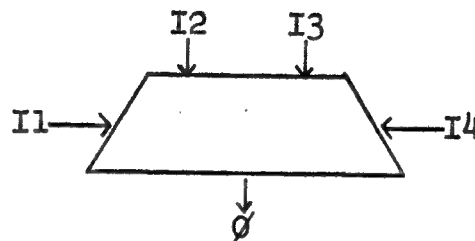
Table 2 (Page 4)

Normally $I1 = 2e^{-BT} \cos WT$

$$I3 = e^{-2BT}$$

where $2B$ is the bandwidth of the filter in radians/sec, W is the center frequency in radians/sec, and T is the sampling interval (1/10000 sec).

14. LINEN I1,I2,I3,I4



This generator produces linear attack and decay envelopes for notes where the attack and decay times are independent of the note duration.

I1 rise time in samples

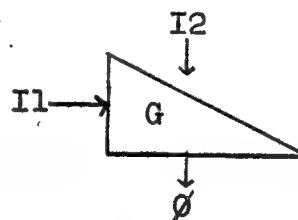
I2 amplitude of the steady state output

I3 duration of entire note in seconds

I4 decay time in samples

$10000I3 - I4$ must be less than 32768. The envelope will not reach full height if $10000I3 < I1 + I4$. If $10000I3 < I1$, then the decay time will be zero.

15. EXPEN I1,I2,G



This unit is most frequently used to generate exponential envelopes according to the equation

$$\phi_1 = I1_1 * G([S_1]_{\text{TRUNCATED}})$$

$$S_{i+1} = S_i + I2_i$$

S_0 is set to zero at the beginning of each note where G is an exponential function. $[S_1]_{\text{TRUNCATED}}$ holds at 512 for all $S_1 > 512$.

EXPEN can also be used for other functions which are to terminate rather than repeat.

Section 6. Music IV routines.

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- 6.1 PASSI
- 6.2 DREAD
- 6.3 WRITER
- 6.4 CON
- 6.5 PASSII
- 6.6 SORT
- 6.7 PASSIII
- 6.8 Orchestra: INITL, GEN, INSERT, PLAY, REMOVE, TER
- 6.9 Output: SPOUT1, SPOUT, SPEND
- 6.10 CVT
- 6.11 GEN07
- 6.12 GEN09
- 6.13 SYSTEM
- 6.14 ENDJOB

6.1 PASSI (main program)

Description - This program reads the score cards which the composer prepares and writes the information derived from these score cards on an intermediate tape, A4, which is subsequently processed in Pass II.

Subroutines which must be provided

DREAD - described in Section 6

WRITER - described in Section 6

ENDJOB - described in Section 6

SYSTEM - described in Section 6

Use of Common Storage

COMMON T,P,G

DIMENSION T(200),P(20),G(12,800)

Details of Operation

The format for the score cards consists of an $\emptyset P$ code plus 12 fields for parameters, P1 through P12, arranged in the card columns as shown:

$\emptyset P$	P1	P2	P3		P12
1-3	4-6	7-12	13-18	etc.	67-72

On cards which directly play notes, P1 is the instrument number; P2 is the starting time of the note measured from the beginning of each section; P3 is the duration of the note. Both P2 and P3 are written in arbitrary time units called beats and the equivalence between beats and seconds is made in Pass II.

Two types of memory are provided in Pass I. The first consists of memory for 800 note card images G(12,800). Entries are made in this memory by putting a number N (1 through 800) in the $\emptyset P$ code field of the note card. The numbers P1 through P12 then are stored in G(1,N) through G(12,N) respectively. A second memory consists of 200 numbers T(200). The parameters T(101) through T(200) are initialized to a value of 1.0 at the beginning of the composition. The parameters T(1) through T(100) are initialized to a value of 0 at the beginning of each section. Any T parameter may be set by means of a card,

ØP	P1	P2
SPF	n	m

where n is the parameter (1 through 200) and m is the value to which it will be set.

A storage space P(20) is provided in common for communication with subroutines. The note card images are read into this space by DREAD. The parameters P1 through P12 go into locations P(1) through P(12). A numerical equivalent of the ØP code goes into P(13). The equivalence table is as follows:

<u>ØP Code</u>	<u>Numerical Equivalence</u>	<u>Meaning</u>
bbb*	1	Play a Note in Pass III
GEN	4	Generate Function in Pass III
SPF	5	Set Parameter in Pass I
TER	6	Terminate Composition
PLF	8	Call Subroutine in Pass I
SEC	9	Terminate Section
SPS	10	Set Parameter in Pass II
ETC	11	Continuation Card
PLS	12	Call Subroutine in Pass II
n	13	Store a Note Card in Pass I or Pass II. n is integer between 1 and 999.

If P(13) = 13, then the number N of the G(12,N) storage array is stored in P(14).

The P array is also used to write data on intermediate tape A4 to communicate with Pass II. For details see the subroutine WRITER. There is a general correspondence between the parameters written on A4 and the parameters written by DREAD into P. P(1) - P(3) are instrument number, starting beat, and duration. P(13) is a numerical ØP code. For GEN cards, P(2) is not the starting time, but rather the starting time in beats is put in P(14).

*bbb stands for a blank ØP code field.

WRITER also takes care of parameters T(1) through T(32). T(1) through T(30) contain the termination times of the note with the latest termination that has been played on instruments 1 through 30 respectively. T(31) contains the termination time of the latest terminating note already played in the section. T(32) contains a count of the number of notes already played. A maximum of a thousand notes may be played in a section and WRITER will terminate the composition with a nasty comment if this number is exceeded.

The operation of the various \emptyset P codes will next be described even though they do not all operate in Pass I.

bbb - Play a note in Pass III

bbb P1 P2 P3 P4...P12

This card causes a note to be played on an instrument P1 starting at time P2 and lasting for duration P3. A note card image is generated on intermediate tapes A4 and B2.

GEN - Generate function in Pass III

GEN P1 P2 P3...P12

This causes the generation of a function in Pass III with subroutine P1. The function number is P2. The subroutine may use parameters P3 through P12. If additional parameters are required they may be written on ETC cards following the GEN card. The function is generated at the time in Pass III equal to the value of T(31) in Pass I when the GEN card was read.

SPF - Set parameter in Pass I

SPF P1 P2

This card causes T(P1) to be set to the value P2 in Pass I.

TER - Terminate composition

This card terminates the composition. In Pass I it terminates the first pass and transfers control by CALL SYSTEM; in Pass II it terminates the second pass and transfers control by CALL SYSTEM; in Pass III it terminates the acoustic output tape and transfers control by CALL SYSTEM. It is also equivalent to an SEC card for the last section in the composition.

PLF - Call subroutine in Pass I

PLF P1 P2...P12

This card calls upon subroutine PLFP1 in Pass I where P1 is an integer, 1 to 15. At the subroutine execution time the parameters P1 through P12 are in the P(20) array. The PLF subroutines which are called upon must, of course, be supplied by the programmer.

SEC - Terminate section

This card terminates a section, resets the timeclock of the compiler and reinitializes T(1) through T(100) to 0. Thus, time which is written in P2 of the note cards is always measured from the beginning of the current section. No more than a thousand notes may be played in a section. This limit is imposed by the sorting in Pass II which is done a section at a time.

SPS - Set parameter in Pass II

SPS P1 P2

This card sets parameter T(P1) to the value P2 in Pass II.

ETC - Continuation card

ETC P1...P12

This card is used for continuing the number of parameters beyond 12. The parameters on the ETC card are always carried along with the previous card. In Pass III the P array is increased to P(480) and the parameters are put in successive blocks of 12 locations in this augmented array. Thus, the 12 parameters on the first ETC card would go into P(13) through P(24), the next ETC card into P(25) through P(36), etc.

PLS - Call subroutine in Pass II

PLS P1 P2...P12

This card calls upon subroutine PLSPl in Pass II where P1 is an integer, 1 to 15. The quantities P2 through P12 will not be in the P(20) array at the time the subroutine is called but they can be referred to by the subroutine. For details read the Pass II FORTRAN program. The subroutine is executed at the end of the section after the note cards have been sorted into chronological order.

n - Store a note card in Pass I or Pass II

n P1...P12

where n is integer, 1 to 999

If n is equal to or less than 800, this card results in the storage of the parameters P1 through P12 in the note card storage space G(1,n) through G(12,n). If n is 801 through 999, this card results in the storage of parameters in the Pass II G array in locations G(12,n-800). Thus, provision has been made for 800 note card images to be stored in Pass I and 199 in Pass II. The storage of a note card image does not directly result in playing a note in Pass III. In order to produce a note one of the PLF subroutines must generate a note card image on tape A4 or B2.

6.10 CVT n Function - Note this is a FORTRAN function subprogram.

Description - The CVT n functions may be called upon by the orchestra program at the beginning of each note. They convert the parameters from the score card into an input parameter for some unit generator. Function CVT n is called by a statement containing Cn in the orchestra definition. All the CVT routines are written by the user of the music program, and we give here only the form of them.

Subroutines which must be provided

Depends on user.

Use of Common Storage

COMMON P

DIMENSION P(480)

Details of Operation

When CVT is called, the P array contains the parameters P1-P12 from the score card in locations P(1) through P(12). The parameters from any following ETC cards are in P(13)-P(24), P(25)-P(36), etc. The CVT function may use these values in computing its value.

6.11 GENO7 Subroutine

Description - This program generates a stored function out of straight line segments.

Calling Sequence

CALL GENO7 (A)

A is an array of dimension 512 in which the output is stored.

Subroutines which must be provided

NONE

Use of Common Storage

COMMON P

DIMENSION P(480)

Details of Operation

GENO7 computes $A(I)$ for $I=1,512$ as points on the line segment function described by the P array. The P array designation is as follows

P(3) - A_1		
P(4) - N_1	P(16) - N_5	P(28) - N_9
P(5) - A_2	P(17) - A_6	P(29) - A_{10}
P(6) - N_2	P(18) - N_6	etc.
P(7) - A_3	P(19) - A_7	
P(8) - N_3	P(20) - N_7	
P(9) - A_4	P(21) - A_8	
P(10) - N_4	P(22) - N_8	
P(11) - A_5	P(23) - A_9	

where A_1 is the amplitude of point P_1 , N_1 is the abscissa distance from point P_1 to point P_2 , A_2 is the amplitude of point P_2 , N_2 is the abscissa distance between point P_2 and P_3 , etc. The line segment function is formed by connecting points P_1 to P_2 , P_2 to P_3 , etc. with straight lines. GENO7 terminates the function when either $N_1 = 0$ or when the sum of the N_1 's is greater than 512.

6.12 GEN09 Subroutine

Description - This program generates a stored function as the sum of segments of sinusoids.

Calling Sequence

CALL GEN09 (A)

A is an array of dimension 512 in which the output is stored.

Subroutines which must be provided

GEN09 is written in FORTRAN and requires a sine function

SINDF(X)

which produces the sine of an argument given in degrees.

Use of Common Storage

COMMON P

DIMENSION P(480)

Details of Operation

The parameters of the sinusoids are given in the P array in the following order

P(3) - A ₁	P(15) - A ₃	etc.
P(4) - H ₁	P(16) - H ₃	
P(5) - P ₁	P(17) - P ₃	
P(6) - B ₁	P(18) - B ₃	
P(7) - E ₁	P(19) - E ₃	
P(8) - A ₂	P(20) - etc.	
P(9) - H ₂	P(21) -	
P(10) - P ₂	P(22) -	
P(11) - B ₂	P(23) -	
P(12) - E ₂	P(24) -	

The function is computed according to the equations

$$A(I) = M \cdot \sum F_j(I)$$

$$F_j(I) = \begin{cases} A_j \text{ SINFD } \left\{ \frac{360}{512} \left[(I-B_j) * H_j + P_j \right] \right\} \\ \text{for } B_j \leq I \leq |E_j| \\ 0 \quad \text{otherwise} \end{cases}$$

The summation is terminated at the first zero value of A_j . If the first E_1 is negative on the last card used to specify the function, then the scaling constant M is set equal to one. If this E_1 is positive, the scaling constant is set so that the maximum absolute value of the function is .9999.